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**REPORT NO.**

7.4

**TITLE:** MACHINE CODED SPEECH

**AUTHOR:** John Schill

**DATE:** 1 December 1968

**SPONSOR:** AFOSR 7-67, AF-3890 NASA NGR

**BIOLOGICAL COMPUTER LABORATORY**

DEPARTMENT OF ELECTRICAL ENGINEERING, UNIVERSITY OF ILLINOIS, URBANA, ILLINOIS

AUTOMATIC DETERMINATION OF INVARIANCE  
IN MACHINE CODED SPEECH

by

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BCL Report 7.4

December 1, 1968

This work was jointly sponsored by

U.S. Air Force Office of Scientific Research  
AF-AFOSR 7-67

U. S. Air Force Systems Engineering Group  
AF 33(615)-3890

National Aeronautics and Space Administration  
Grant NGR 14-005-111

BIOLOGICAL COMPUTER LABORATORY  
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URBANA, ILLINOIS



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## 1. INTRODUCTION

### 1.1 General Remarks

One of the major problems in speech recognition is the excessive complexity of the signal that is to be analyzed. Each syllable of a spoken word may involve several motions of the articulatory tract to generate the desired sound. Since each of these motions must produce a different sound, a speech analyzer must produce a corresponding differentiation in its output for each of these different sounds. Hence much data are produced for even a simple sentence. A method for handling these data conveniently and for presenting them efficiently was necessary. With the collection of these data made, a method of analysis to determine uniquely the invariant patterns for each spoken word had to be found. Such a method for data collection and analysis of speech will be described in the following pages.

### 1.2 Analysis and Synthesis: Early Developments

Speech is usually accepted as the most natural method for the transmission of information for man, and the history of "talking automata" goes very far back, indeed. "Speaking machines" developed in the eighteenth century can still be seen in museums. The most remarkable one was built by

Wolfgang von Kempelen.<sup>5</sup> His automaton utilized reeds to excite hand varied resonators to produce voiced sounds, while bellows provided air to the reeds. Restricted passages controlled by the operator produced the consonants. Von Kempelen's machine utilized the basic principle of modern vocoders, i.e., separating the "voicing sound generator" with little information from a "controller" with almost all the information.

The first systematic study of speech sounds, however, was done by Helmholtz in the nineteenth century.<sup>8</sup> He was able to synthesize vowel sounds by vibrating a number of pitchforks at selected frequencies. The first recording of the waveforms of speech sounds was made by a device called the "phonautograph." Up to the early twenties, many investigators of speech suspected that speech sounds are characterized by multiple resonances. At that time, Sir Richard Paget<sup>13</sup> determined that vowels and other sounds were characterized by two or three dominant resonances in the vocal tract.

### 1.3 Analysis and Synthesis: Recent Developments

After these promising beginnings, speech in the last few decades has been the subject of intensive research effort as a result of recent developments in the theory of



communication and artificial intelligence. The goal of this machine-oriented speech research is an effective man-machine verbal communications system. This system must convert speech into a code that permits machine recognition of speech, and ultimately must be able to synthesize speech. The first step toward this goal was Dudley's "vocoder."<sup>5</sup>

A vocoder is a bandwidth compression system which analyzes and later resynthesizes speech from transmitted control signals. Gazdag says, "The basic idea, from an articulatory point of view, is to transmit only signals to specify the movements of the vocal tract for the purpose of controlling an artificial vocal tract at the receiving end."<sup>7</sup>

The first vocoder utilized a set of bandpass filters of a bandwidth of 150 Hz. The rectified and smoothed filter outputs were modulated to synthesize the sound. A summation of the modulated control signals formed the resynthesized speech which was fairly intelligible but lacked naturalness. It was found that the transmission of the control signals required only 10 per cent of the bandwidth covered by the filters which in theory meant that only 10 per cent of the channel capacity is required to transmit the speech signals.

Since the first vocoder, there has been, despite many efforts, no major breakthrough in vocoding techniques. There have been, however, improvements in circuitry and components, but the basic principles have not appreciably changed.

#### 1.4 Short-time Spectrum

The area of speech recognition is quite allied to that of the vocoders as the output of the filters of the vocoder do yield enough information for one to be able to understand what is being spoken into the filters. The collection of all the instantaneous filter outputs is referred to as the "short-time spectrum." Visually displayed, these spectra are clearly distinguishable and can be read with reasonable accuracy (see Figure 1 for example).

It has been noted that the "short-time spectrum" has peaks at various frequencies, known as "formant frequencies." These peaks seem to occur more or less periodically. The set of peaks at the lowest frequencies is known as the first formant; the next lowest, the second, etc. In 1952, using the first three formants, David Biddulph and Balashek<sup>3</sup> were able to recognize the spoken digits, "one," "two," ..., "ten," with nearly 98 per cent accuracy for a system adjusted to the speaker's voice.

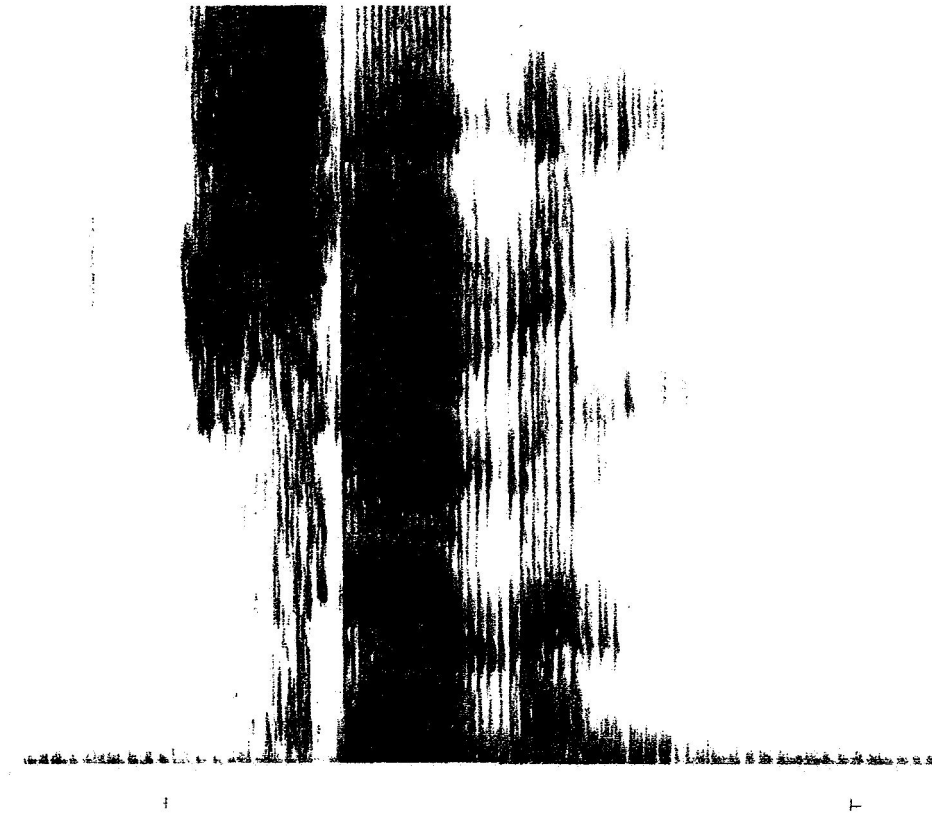
**TYPE B SONAGRAM ®**

Figure 1. A Sonagram for the Word "SEVEN"

There have been many other systems since this, but all were limited to the recognition of vowels in the well defined context of the ten spoken digits pronounced under exact laboratory conditions.

The difficulties in speech recognition are quite significant. Speech by its very nature is not a "nice" continuous signal amenable to the basic tools of an engineer. Everyone speaks differently; moreover, each person's voice changes with various situations. The code of speech is not simple to break.

#### 1.5 Gazdag's Decoder

Gazdag<sup>7</sup> has attacked this problem in a rather interesting manner. He challenged the usefulness of phonetic concepts as, e.g., "phonemes" in speech analysis based on mechanical processors. Instead of using phonemes as the basic speech units, he suggested letting the design of the analyzing system define these basic units which he called "machine events." Machine events are in his specific case a function of the set of output signals of the filters of a well defined system he called "Processor." Some phonemes may be machine events but others may be a series of machine events. The definition of a machine event is characterized

by the structure and operation of the machine used. Machine events are functionally defined. They yield information about the articulation of the speaker in order to differentiate conveniently his speech sounds.

Gazdag gives two requirements for such a Processor. The Processor must operate invariantly with respect to various intensities of like utterances; secondly, it must operate invariantly with respect to temporal variations of like utterances. The reasons for these requirements are fairly obvious. Intensity invariance is needed because one usually does not speak at the same volume constantly; similarly, invariance with respect to time variation is required since one does not always speak at the same rate.

Let us consider the Processor as a part of a larger system--the Decoder. When a written message is read by a speaker into the input of Decoder, a reproduction of the written message appears at the output of the Decoder. In a functional sense, the Decoder can be thought of as a system that performs some kind of a many-to-one mapping from the signal space to the message space. The system's input is a continuously varying speech signal, while its output can assume only a finite number of discrete states. The message is expressed in terms of these output states.

Two major operations are performed by the Decoder (Figure 2). The first stage is called the "Processor." It transforms the speech signal into machine events as described above. The second stage is called the "Translator." This stage operates on the output of the "Processor," translating sequences of machine events into output messages.

First, let us consider the Processor. It has only one input channel but has  $n$  binary output channels,

$$u_i; i=1, \dots, n$$

where the values that the channel's signal  $u_i$  may assume are

$$u_i = 0, 1$$

At any instant, the set of outputs of the Processor will be interpreted as a row vector.

$$U = (u_1, u_2, \dots, u_n)$$

which can assume  $2^n$  possible states. Each of these states signifies a machine event which in turn may be thought of as the machine representation of a set of articulatory positions of a speaker's vocal tract. At this point, it is necessary to postulate that when a speaker generates a word, his articulators move simultaneously and continuously to produce a sequence of articulatory positions. Therefore, when the speaker utters a word (or part of it) at slower or faster rates, he executes approximately the same sequence

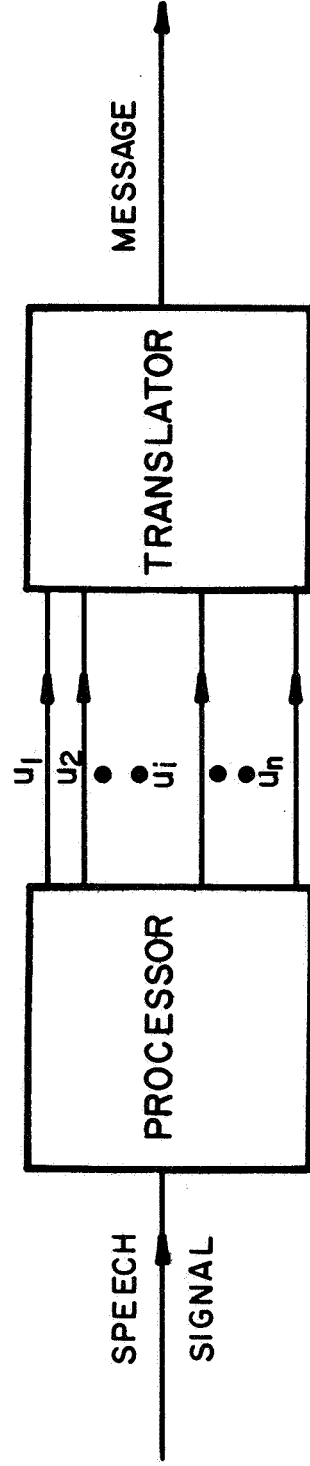


Figure 2. Gazdag's Decoder.

of articulatory gestures at a slower or faster rate. The necessity of this postulate is obvious from the viewpoint of parsimony as well as common sense. Even if we spoke using different articulatory positions at different rates while saying the same word, the sequence of sounds we hear must have considerable similarity; otherwise the different utterances would not sound as though they represented the same word. Hence the information about the identity of a spoken word is represented by the order of occurrence of a sequence of  $N$  machine events.

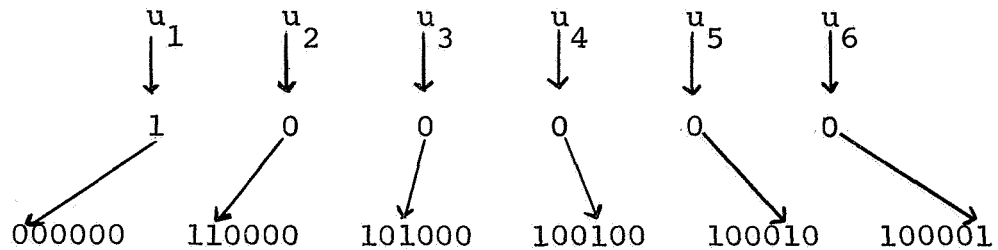
$$U_0, U_1, U_2, \dots, U_N$$

The duration of each  $U_j$  provides an indication of the rate of utterance but its duration is irrelevant as printed equivalent of the utterance.

Each  $u_i$  of the row vector represents independent properties relating to the utterances; therefore if independent in value, the probability of more than one element of the row vector changing at the same instant is infinitesimally small for practical purposes. Therefore the Hamming distance between two adjacent machine events must be unity. This means that only one component,  $u_i$ , of two adjacent row vectors changes its value. For example, consider the



following row vector:



It can change only to the row vectors shown above.

Let us consider what this implies for the decoding of speech. First, consider the sequence of  $N$  machine events,

$$U_0, U_1, U_2, \dots, U_N$$

where  $U_i$  is the  $n$ -digit binary number

$$u_{i,1}, u_{i,2}, \dots, u_{i,n}$$

Since the possible representation of each machine event is  $2^n$ , the number of possible representations of a spoken word of a sequence of  $N$  machine events is  $2^{nN}$  if the constraint of a Hamming distance of one unit for adjacent machine events does not hold. For example, the spoken digits, "one," "two," "three," etc., are represented by between 8 and 15 machine events, while more polysyllabic words will, of course, require many more machine events to represent them. This means that if the machine representation of a spoken word was a sequence, say of 10 machine events ( $N=10$ ),

each with 6 output channels ( $n=6$ ), there could be  $2^{60}$  or approximately  $10^{18}$  possible representations. However, taking the constraint of a Hamming distance of one unit only between subsequent machine events into consideration, the number of possible representations decreases drastically. For the representation of 10 machine events with 6 output channels as before, there are only  $1.1 \times 10^{10}$ , or approximately  $2^{34}$  possible representations. In other words, the constraint on two subsequent machine events provides a code that reduces the entropy of unconstrained sequences of 60 bits, as in the example above, to slightly more than 34 bits.

Because of the ambiguity of the encoding process, a set of speech events for a particular spoken word by a particular speaker may vary from trial to trial. However, due to the fact that the vocal tract goes through approximately the same process for each utterance of a word, the sequences should possess many of the same machine events. Gazdag calls the sequence of common machine events a "Significant Subsequence," (SSS). This means that in repeated utterances of the same word, W

$$S_1(W); U_{1,0}, U_{1,1}, U_{1,2}, \dots, U_{1,j}, \dots, U_{1,N_1}$$

$$S_i(W); U_{i,0}, U_{i,1}, \dots, U_{i,j}, \dots, U_{i,N_i}$$

$$S_r(W); U_{r,0}, U_{r,1}, \dots, U_{r,j}, \dots, U_{r,N_r}$$

(where  $S_i(W)$  represents the sequence of  $N_i$  machine events of the  $i$ -th utterance of word and  $W_{U_{k,l}}$  represents in the  $k$ th utterance of the same word  $W$  the  $l$ th machine event in the representation of that utterance), there exists the largest common sequence of speech events that is sufficient to signify this particular word.

Let us consider the sequence  $S_i(W)$ . It is an ordered set of  $N_i$  elements. The number of ordered subsets of  $m$  elements for  $S_i(W)$  is

$$\mathcal{N}(N_i, m) = \binom{N_i}{m}$$

The total number of non-empty subsets is

$$\mathcal{N}(N) = 2^{N_i} - 1$$

For each utterance  $S_i(W)$ , all the subsets are established:

$\mathcal{N}(N_i)$  (i.e., the set of all ordered subsets). We now

form the intersection of all the  $SS(N_i)$

$$\bigcap_i SS(N_i)$$

and select the elements with largest  $m$  of this new set. If there is one and only one, this is the SSS. If there are more, we iterate.

Let us look at some of the characteristics of these SSS's. Consider the following SSS for the word  $W$  of repeated utterances.

$$U_{1}^{*}, U_{2}^{*}, \dots, U_{j}^{*}, \dots, U_{N}^{*}$$

where  $U_{j}^{*}$  represents the  $j$ th machine event of the SSS for the word  $W$ . The most obvious characteristic is that each machine event,  $U_{j}^{*}$ , is included in each  $S_i(W)$ ; otherwise it could not be a characteristic part of that word  $W$ . If this were not the case, the sound of the vowel "u" could be said to be characteristic of the sound of the word "car."

The second characteristic is the sequential nature of the subsequence itself. The particular sound represented by  $U_{i-1}^{*}$  must be followed by, but does not necessarily have to be, immediately adjacent to,  $U_i^{*}$  in a particular original sequence. For example,  $U_8^{*}$  may correspond to the tenth machine event,  $U_{i,10}$ , in a particular sequence  $i$ , while  $U_9^{*}$  may correspond to the fourteenth machine event,  $U_{i,14}$ . These two characteristics are the two major pillars of the hypothesis which claims that a spoken word can be identified by its SSS.

Let us now briefly consider the Processor used in this experiment, for the specific details are discussed elsewhere.<sup>7</sup> (See Figure 3.) In general, this Processor consists of twelve filters which cover the frequency spectrum from 150 Hz to 3000 Hz. Each covers a particular band of this spectrum. The outputs of the filters,  $a_i$ , are analog in nature, thereby yielding information not only of the existence of a component at that filter's frequency but also its magnitude.

The information content of the set of these filter outputs is redundant<sup>15</sup>, and the first problem is to find a way to reduce this redundancy. If we consider the set of all filter outputs to be represented by a twelve dimensional space, the instantaneous state of all filters may be represented by a vector in this space. After considerable experimentation, Gazdag was able to show that the essential information about speech sounds is preserved by locating this vector in well defined regions of this space that are separated by six strategically placed hyperplanes. These hyperplanes partition the twelve dimensional space into 64 regions. The vector in this twelve dimensional space which represents the particular sound entering the filters at a given instant in time moves through these 64 regions,

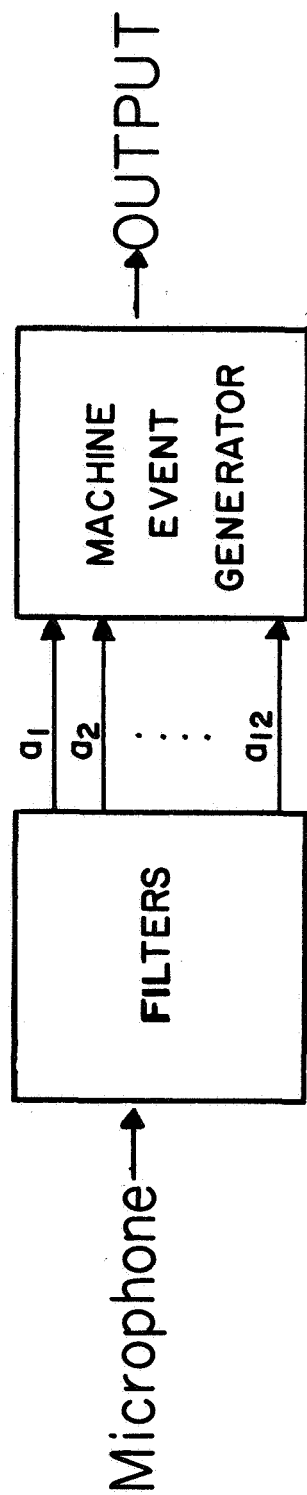


Figure 3. The Processor

causing the machine event generator (Figure 3) to have a specific output state corresponding to the region in which the vector is at that given instant. These output states are called the "machine events" which were discussed earlier. Due to the time response of the filter outputs, any machine event to be considered meaningful must persist for at least 20 msec. Using this criterion plus the one unit Hamming distance constraint, Gazdag was able to determine the SSS for a specific word, given several strip recordings from the machine event generator of repeated utterances of that word. (See Figure 4 for examples.) This procedure is tedious and time-consuming. The problem that will be dealt with in the following pages is how to define operationally this procedure of determination of these SSS's and how to assemble a system to do this determination automatically.

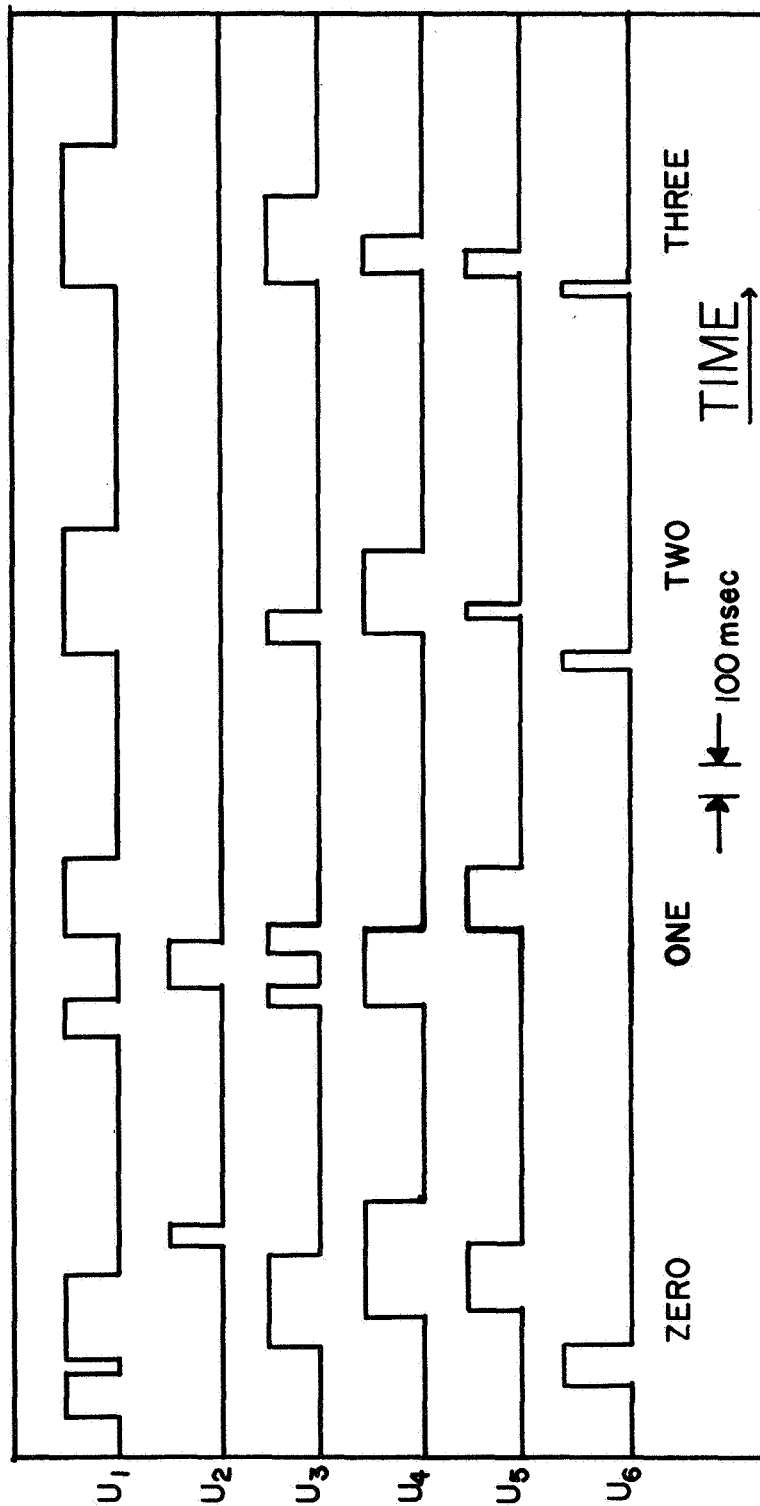


Figure 4. Examples of the Processor's Output.



## 2. THE ALGORITHM

Before any hardware can be made, it is necessary to find first an algorithm that computes significant sub-sequences, (SSS). Let us consider Table 1. Each column represents the machine encoding of a particular spoken word "W"--in this case it is the digit "ONE"--in the form of a sequence of machine events. In developing the desired algorithm, one sequence may be chosen as the standard of comparison to which the other sequences are then compared.

Let us pick the sequence  $S_1$  as the standard. The first member of the standard,  $U_{1,1}$ , is compared with the members of the other sequences in the following manner. When the  $S_2$  sequence for the comparison is used,  $U_{1,1}$  is compared to  $U_{2,1}$ . If it is identical, the fact is noted and the next sequence, say  $S_3$ , is inspected. If  $U_{1,1}$  and  $U_{2,1}$  are not identical  $U_{1,1}$  is compared to  $U_{2,2}$ . If they are identical, the fact is noted and the next sequence is inspected. If  $U_{1,1}$  and  $U_{2,2}$  are identical, the process is continued until a match is found or a quarter of the length of the  $S_2$  (the LOWER SEARCH LIMIT) has been searched. If no match has been found, the fact is noted and the next sequence is searched in the same way as the  $S_2$  sequence was

$S_1(W)$	$S_2(W)$	$S_3(W)$	$S_4(W)$	$S_5(W)$
$U_{1,1} = 000000$	$U_{2,1} = 000000$	$U_{3,1} = 000000$	$U_{4,1} = 000000$	$U_{5,1} = 000000$
$U_{1,2} = 100000$	100000	100000	100000	100000
101000	101000	101000	101000	101000
001000	001000	001000	001000	101100
001100	001100	000000	000000	100100
000100	000100	000100	000100	000100
010100	010100	010100	010100	010100
010110	010110	010110	010110	010110
011110	011110	011110	011110	011110
111110	111110	111110	111110	111110
101110	101110	101110	101110	101110
101010	101010	101010	101010	101010
100010	100010	100010	100010	101110
100000	100000	100000	100000	101010
$U_{1,15} = 000000$	$U_{2,15} = 000000$	$U_{3,15} = 000000$	$U_{4,15} = 000000$	100010
				100000
				$U_{5,17} = 000000$

Table 1. Examples of the Machine Events  
for Repeated Utterances of the Word "ONE"

inspected. The process continues until the sequences other than the chosen sequence have been inspected for the existence of a machine event identical to  $U_{1,1}$ . Once all sequences have been searched, a tally--or vote--of the number of sequences having a match for  $U_{1,1}$  is taken. If all the sequences, or all but one sequence, have a match for  $U_{1,1}$ , then  $U_{1,1}$  is considered a COMMON MACHINE EVENT. This is called the "ALL BUT ONE CONSTRAINT." In all the sequences which have a match for  $U_{1,1}$  the UPPER SEARCH LIMIT is set equal to the sequential location plus one of the match  $U_{1,1}$ . This means that in searching for a match for  $U_{1,2}$ , the search will begin in a particular sequence at the upper search limit and continue until a match is found or a quarter of the length of the sequence beyond the upper search limit or the end of the sequence, whichever comes first, is inspected. In the case of a sequence without a match for  $U_{1,1}$ , the upper search limit is the first machine event. All common elements taken in order of discovery form the SSS.

When the member of the chosen sequence is located toward the end of the sequence, the lower search limit of the other sequence (i.e., the upper search limit plus a quarter of the sequence length) may exceed the length of that

sequence. Therefore it is necessary to redefine the lower search limit as the last member of that sequence. (See Figure 5 for the cases.)

There is also a problem if the last members of two sequences are chosen as common subevents, and there still are elements in the chosen sequence to be tested for matching. In this case, the algorithm is ended since the criterion for a common machine event cannot be satisfied.

There is one weakness with the algorithm as it now stands. Consider the following sequence of numbers representing machine events in sequences  $S_1, S_2, \dots, S_5$

$S_1:$  1 2 3 1 4 7 6 3  
 $S_2:$  1 2 3 4 1 7 6 3  
 $S_3:$  1 2 3 5 1 4 1 7 6 3  
 $S_4:$  1 5 2 3 1 4 1 7 6 3  
 $S_5:$  1 2 3 1 2 4 1 8 7 6 3

If sequence  $S_1$  were used as the chosen standard, the SSS is

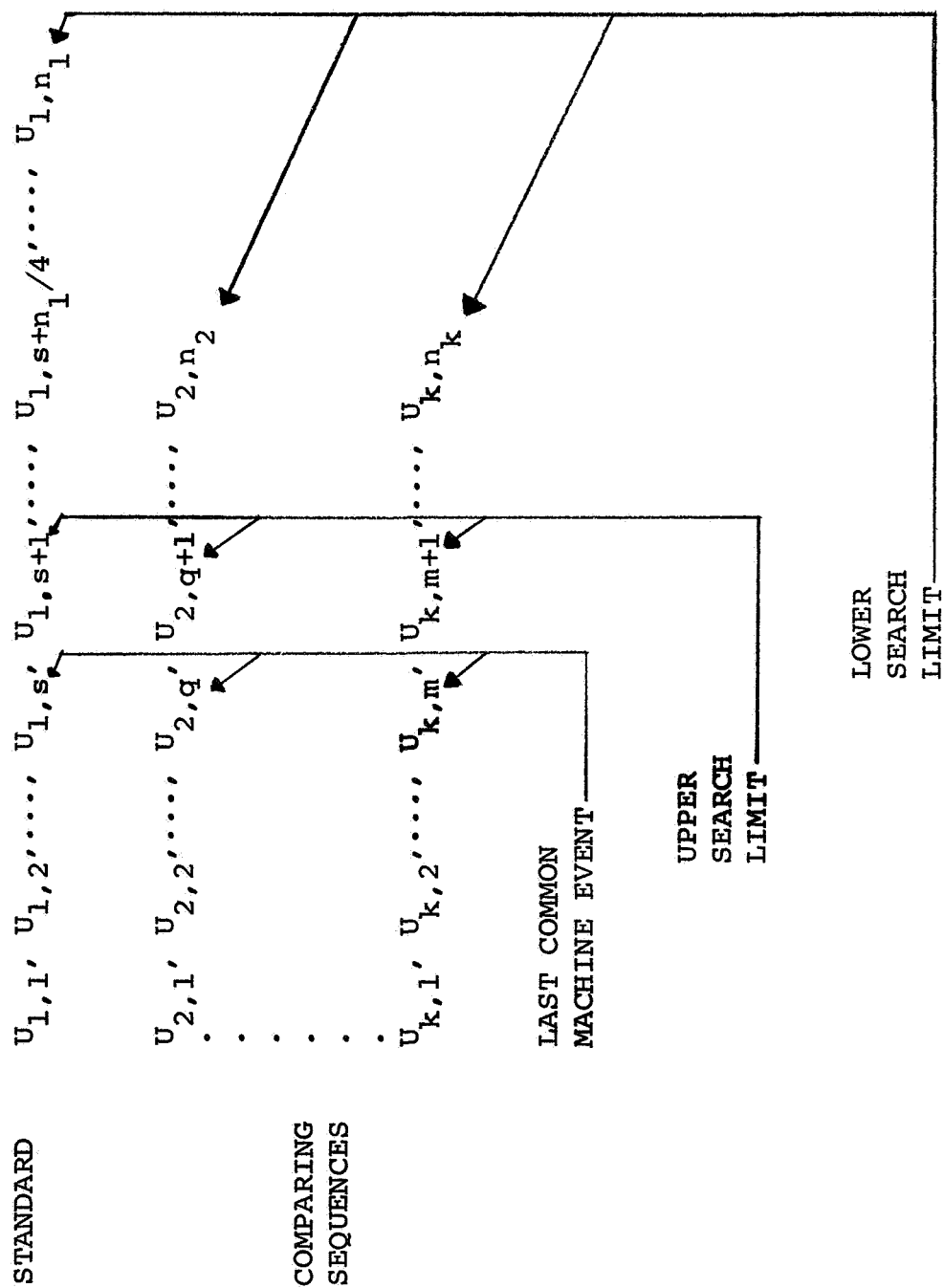
$SSS_1:$  1 2 3 1 4 7 6 3

If sequence  $S_2$  were used as the chosen standard, the SSS is

$SSS_2:$  1 2 3 4 1 7 6 3

If any of the remaining  $S_i$ 's were used as the chosen standard, the SSS is

$SSS_i:$  1 2 3 1 4 1 7 6 3,  $i = 3, 4, 5.$



Where  $U_{a,b}$  is in the  $a$ -th sequence the  $b$ -th machine event

and  $s + n_1/4 < n_1, q + n_2/4 > n_2, m + n_k/4 > n_k$ .

Figure 5. An Example of the Algorithm in Progress

To eliminate this ambiguity, each sequence is used as the standard, the set of SSS's generated is treated as a set of sequences and put through the algorithm again! The result of this iterated application of the algorithm produces uniquely the "True Significant Subsequence,"  $SSS^*$ . For example, using the algorithm on the five sequences above, we find the following:

Using  $S_1$  as the standard, the SSS is

1 2 3 1 4 7 6 3

Using  $S_2$ : 1 2 3 4 1 7 6 3

Using  $S_3$ : 1 2 3 1 4 1 7 6 3

Using  $S_4$ : 1 2 3 1 4 1 7 6 3

Using  $S_5$ : 1 2 3 1 4 1 7 6 3

Now when the set of SSS's (denoted by  $SSS^{(1)}$ ) is again put through the algorithm, using the "ALL BUT ONE CONSTRAINT," the same SSS's are again obtained. In other words,  $SSS^{(1)} = SSS^{(2)}$ .

But since these SSS's are not identical to each other, it is necessary to tighten the "ALL BUT ONE CONSTRAINT" to a constraint for which a machine event in the SSS must occur in all sequences being performed upon by the algorithm. Hence in our example, using this constraint, one obtains a

unique sequence

SSS<sup>\*</sup>: 1 2 3 4 7 6 3

which represents the "true" SSS, or SSS<sup>\*</sup> for this particular utterance.

For a set of sequences of short length in the example as above, it is sufficient to operate the algorithm on these sequences and also their SSS's to find the true SSS, SSS<sup>\*</sup>. For a set of more lengthy sequences, a more complex process is necessary. The original sequences, called the 0th order SSS or SSS<sup>(0)</sup>, are operated on by the algorithm to find a set of SSS<sup>(1)</sup>, called the first order SSS's. If all the SSS's<sup>(1)</sup> are identical, then the true SSS is any of these first order SSS's, i.e., SSS<sup>(1)</sup> = SSS<sup>\*</sup>. If such is not the case, the algorithm is used on the first order SSS to get the second order SSS's, SSS<sup>(2)</sup>. If these are not identical to each other, but identical to SSS<sup>(1)</sup>, then the "ALL BUT ONE CONSTRAINT" is stiffened as in the example and the SSS<sup>(1)</sup> are again operated upon by the algorithm to find the SSS<sup>\*</sup>. If the SSS<sup>(2)</sup>, are identical, the SSS<sup>\*</sup> is found. If neither of these cases holds, the algorithm is used until there is found a set of identical SSS's, the nth order SSS's, which is the true SSS, i.e., SSS<sup>\*</sup>. It can be conjectured that since at each stage some of the non-

common elements are eliminated from the SSS's, the SSS\* must exist. No formal proof of this conjecture will be offered here as it is beyond the scope of this paper.

To utilize this algorithm to analyze the sequences for the spoken digits, a computer program was developed, and examples of the zeroth order are given, along with the computed first and second order SSS's. They are shown in Tables 2, 3 and 4 for convenience in base 10 form by utilizing the computer program instead of in their original binary representation. For all cases tried, the spoken digits, the SSS's converged at most by the second order SSS. The data input and output and the flow chart of this program will be discussed Appendix A and B, respectively, along with a card listing.



## 0th Order SSS for the word "ONE"

```

00 32 36 44 12 04 20 28 12 44 40 42 34 32 00
00 32 36 04 20 28 12 44 40 42 34 32 00
00 32 40 44 12 04 20 28 30 14 12 44 40 32 34 32 00
00 32 36 04 12 04 20 28 12 44 40 32 34 32 00
00 32 36 44 12 04 20 28 12 44 40 42 34 32 00

```

## 1st Order SSS for the word "ONE"

```

00 32 36 12 04 20 28 12 44 40 34 32 00
00 32 36 04 20 28 12 44 40 34 32 00
00 32 36 12 04 20 28 12 44 40 34 32 00
00 32 36 04 20 28 12 44 40 34 32 00
00 32 36 12 04 20 28 12 44 40 34 32 00

```

## 2nd Order SSS for the word "ONE"

```

00 32 36 04 20 28 12 44 40 34 32 00
00 32 36 04 20 28 12 44 40 34 32 00
00 32 36 04 20 28 12 44 40 34 32 00
00 32 36 04 20 28 12 44 40 34 32 00
00 32 36 04 20 28 12 44 40 34 32 00

```

$$SSS^* = SSS^{(2)}$$

Table 2. The Significant Subsequences for the Word "ONE"

0th Order SSS for the word "TWO"

00 01 00 32 40 44 36 38 36 32 00

00 20 00 01 00 32 40 44 36 32 00

00 01 00 32 40 44 36 32 00

00 01 00 32 40 44 36 32 00

00 02 00 01 00 01 00 32 40 44 36 32 00

1st Order SSS for the word "TWO"

00 01 00 32 40 44 36 32 00

00 01 00 32 40 44 36 32 00

00 01 00 32 40 44 36 32 00

00 01 00 32 40 44 36 32 00

00 01 00 32 40 44 36 32 00

$$SSS^* = SSS^{(1)}$$

Table 3. The Significant Subsequences for the Word "TWO"

0th Order SSS for the word "THREE"

```

00 01 00 32 40 42 34 38 36 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 01 00 32 34 42 46 44 40 32 00

```

1st Order SSS for the word "THREE"

```

00 32 34 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00

```

2nd Order SSS for the word "THREE"

```

00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00
00 32 34 42 46 44 40 32 00

```

$$SSS^* = SSS^{(2)}$$

Table 4. The Significant Subsequences for the Word "THREE"

### 3. THE SAMPLER

To connect the Processor to the computer which contains the algorithm, the author found it necessary to provide a "filter" as an interface, called the "Sampler" (second rack on the right of Gazdag's Processor; Figure 6). It eliminates most of the noise generated by the Processor as, for instance, the Processor's transient states. The sampler signals to the computer that the Processor has a valid signal. The computer functions as a storage device as well as the implementation of the algorithm.

The sampler samples the output of the Processor and transmits its states to the computer and operates in two modes: a synchronous and an asynchronous mode. In the synchronous mode the rate at which the Processor's output is sampled is controlled by an external clock pulse (CP). In the asynchronous mode the Processor is sampled continuously. In this mode, a change of the Processor's state causes a change in the sampler's output.

The input of the sampler can be of two forms: inputs compatible with integrated circuit logic levels and those which are not. For the latter, the circuit in Figure 7 converts the input level to levels necessary for operation

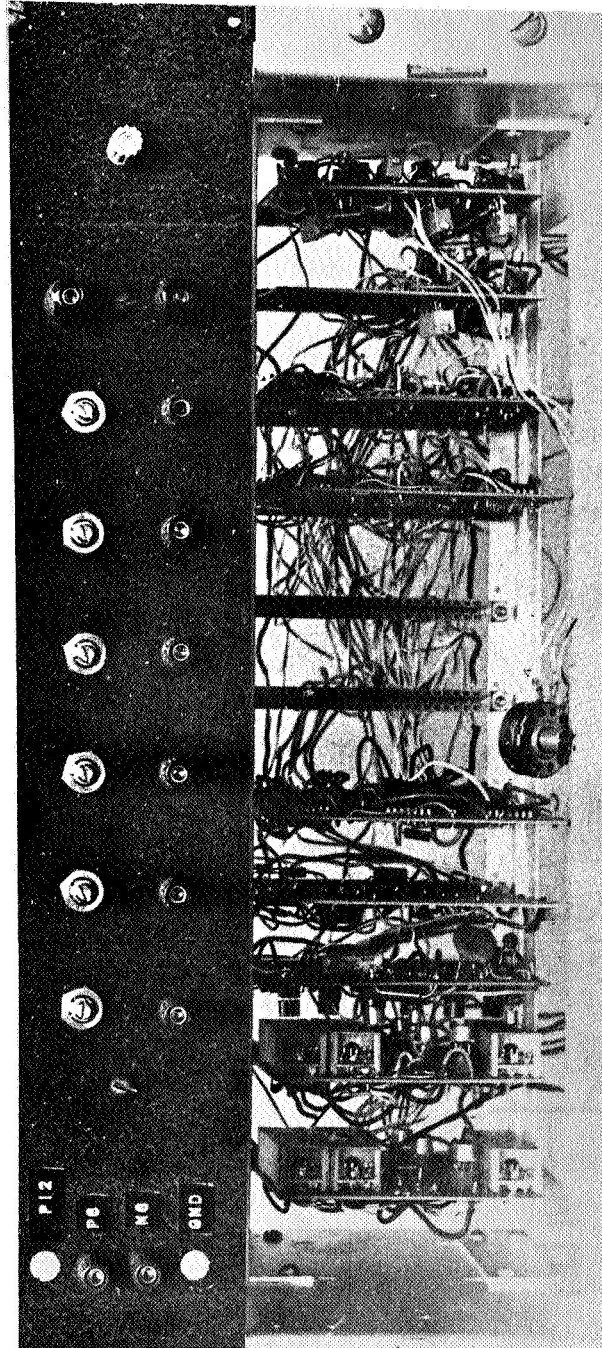


Figure 6. The Processor with the Sampler

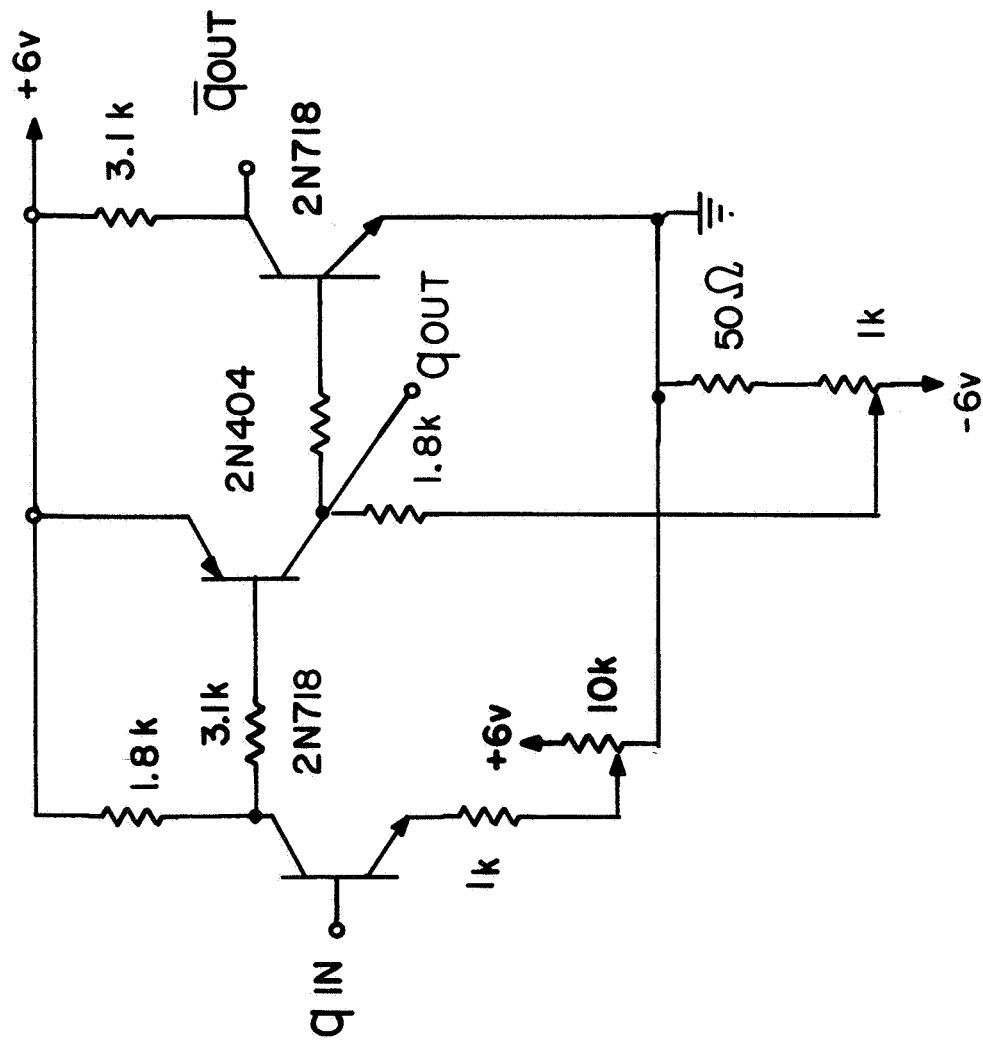


Figure 7. Incompatible Logic Level Converter

of the integrated circuits (+6 volts for high level, zero volts for low level). The potentiometer, R3, controls the input triggering level. The outputs of this circuit give both the input logic level  $q_i$  and its complement,  $\bar{q}_i$ . For compatible inputs the integrated circuit (IC) schematic is shown in Figure 8. The outputs of this circuit are both the input logic level and its complement.

The input logic circuit feeds into comparing circuitry (Figure 9). These inputs feed a group of JK flip-flops operating in a synchronous mode which store the PRESENT STATE,  $q_i$ , of the Processor. Another group of flip-flops in the asynchronous mode contains the LAST STATE,  $q_{l_i}$ , of the Processor. The present and last states are EXCLUSIVE or compared through a series of "NAND" gates. If the two states are different, another asynchronous flip-flop is set into the high state ( $q_i = 1$ ), causing a one millisecond oneshot to activate. The oneshot's output is called the PASS GATE (PG). The PG allows the LAST STATE ( $q_{l_i}$ ) and output flip-flops to change to the PRESENT STATE. (Not shown in Figure 8.) The output flip-flops drive the output circuitry.

The output circuitry (Figure 10) consists of a series of relay drivers with their relays. The drivers enable the

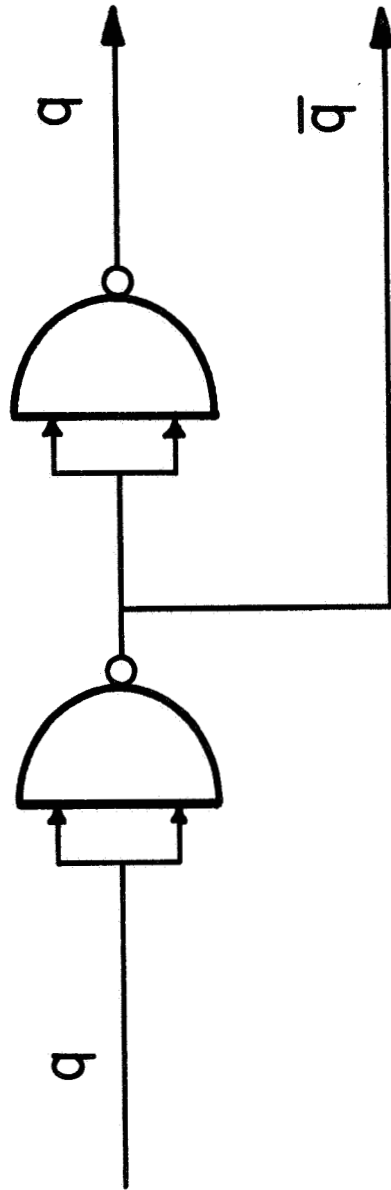


Figure 8. Compatible Logic Level Converter.



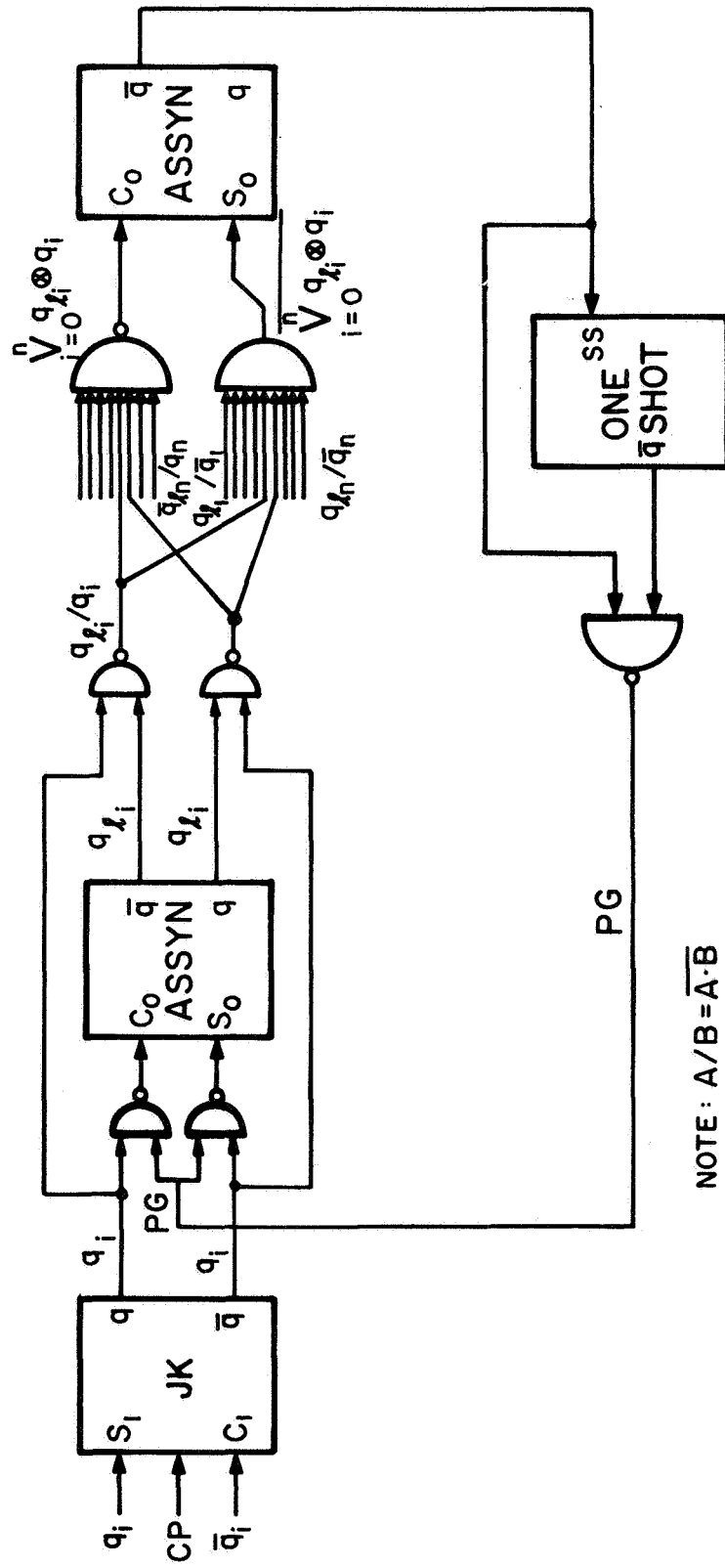


Figure 9. The Comparing Circuitry

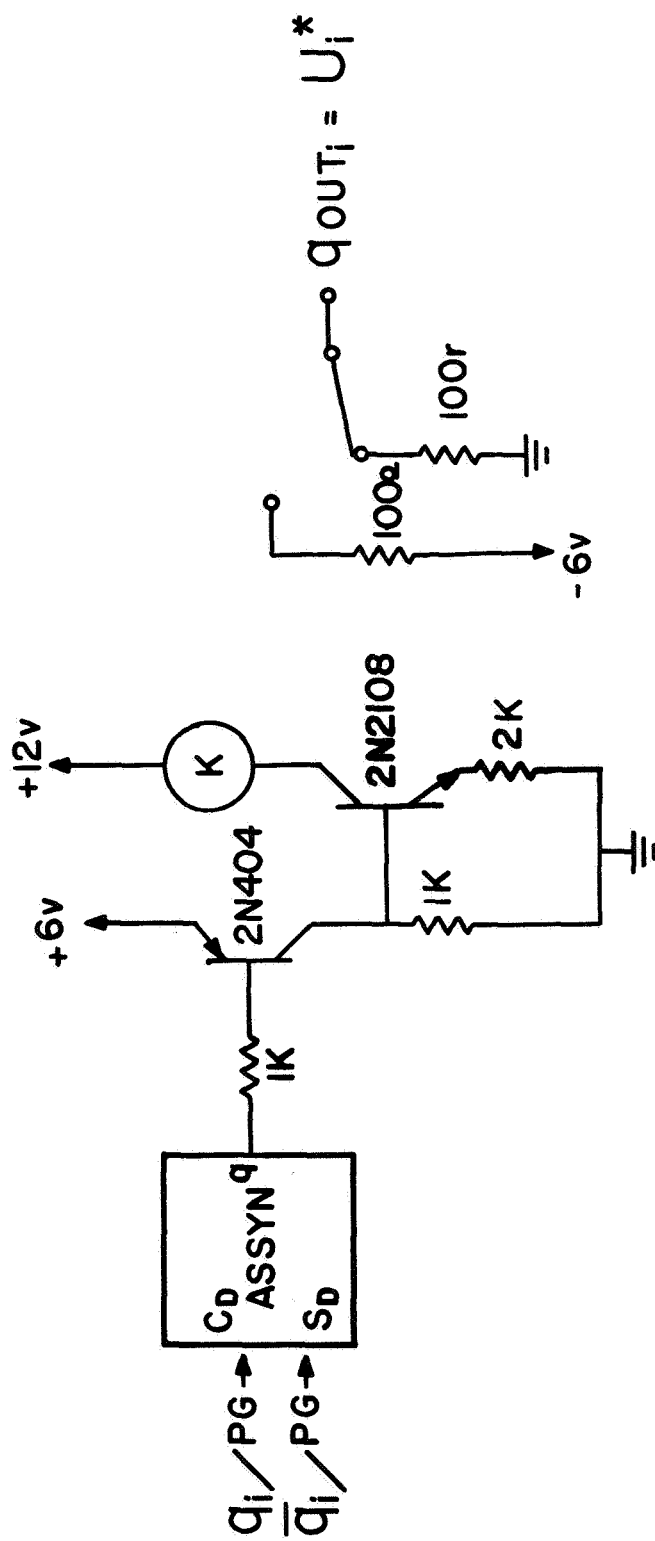


Figure 10. The Output Circuitry

IC logic to drive the relays. From the relay contacts, almost any logic level voltages can be accommodated. For the sampler to be compatible with the Illiac II, it is necessary to have logic levels of 0 (as high value) and -5V (as the low value) with 100 ohms minimum resistance. A 500 nsec oneshot is also included to pulse the Illiac II every time there is a new state (indicated by  $+ = 1$  signal). It should be noted that aside from the level compatibility, the relay response time (about 20 msec) enables the sampler to operate in an asynchronous mode without trouble from transient states.

Figure 11 shows the entire information flow through the speech recognition system. The speech signal enters the microphone of Gazdag's Processor which translates utterances into sequences of machine events. These enter the Sampler at the LOGIC LEVEL CONVERTERS and are stored (sequentially) in the PRESENT STATE flip-flops. By means of the logic circuitry of the Sampler, the valid machine events are passed sequentially to the output relays. From the contacts of these relays, the computer takes the machine events and stores them as the zeroth order SSS to be operated upon by the algorithm program. Finally, the output of the algorithm is in the form of a computer printout. The Sampler and

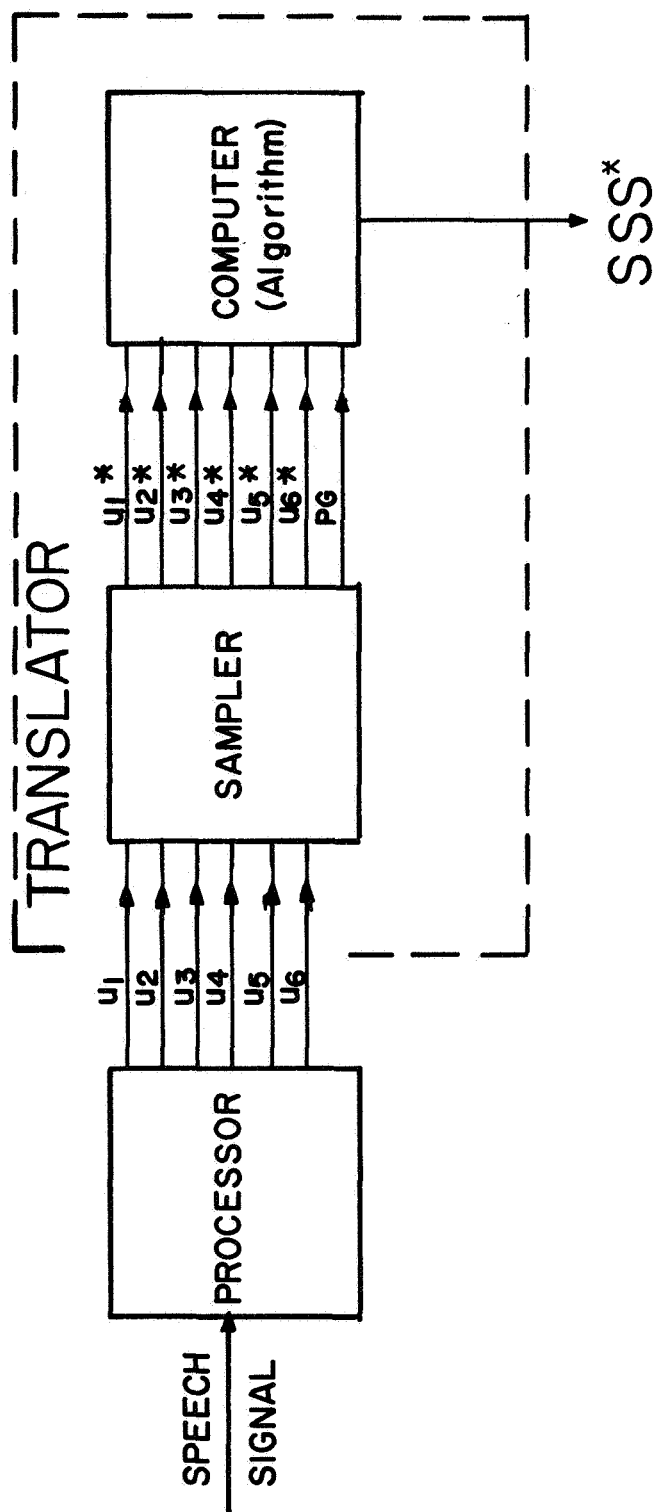


Figure 11. The Speech Recognition System

computer with the algorithm form the Translator of the Decoder. The Translator portion of the Decoder is the subject matter of this paper.

#### 4. CONCLUSION

In this paper, an automatic method for finding invariants (SSS's) in sequences of machine events was presented.

Based on the work of Gazdag, a Sampler for his Processor was built by the author. It passes valid signals, the machine events, from the Processor to a computer. The validity of these machine events is determined by their persistence for more than 20 msec and by a Hamming distance of one unit between the adjacent events. Utilizing this Sampler, a speaker speaking into the Processor is directly connected to the computer which performs the computation of the true significant subsequences, SSS\*. This eliminates the need of manual manipulation of the recorded output of the Processor.

An algorithm was developed to find Gazdag's SSS\*. This algorithm has been implemented on the IBM 7094 and the Illiac II. Several examples of the spoken digits were tested, and indeed a true SSS was found in each case. (See Tables 2, 3 and 4 for some examples.)

With the algorithm and the Sampler it is possible to find the true SSS for words other than the spoken digits. A library of such representations of spoken words can now

be built, making it possible to achieve real-time recognition of human speech.

Although the system described here is still in a relatively simple stage, it may represent a step forward--however small--in the direction of real-time speech recognition and man-machine communication.

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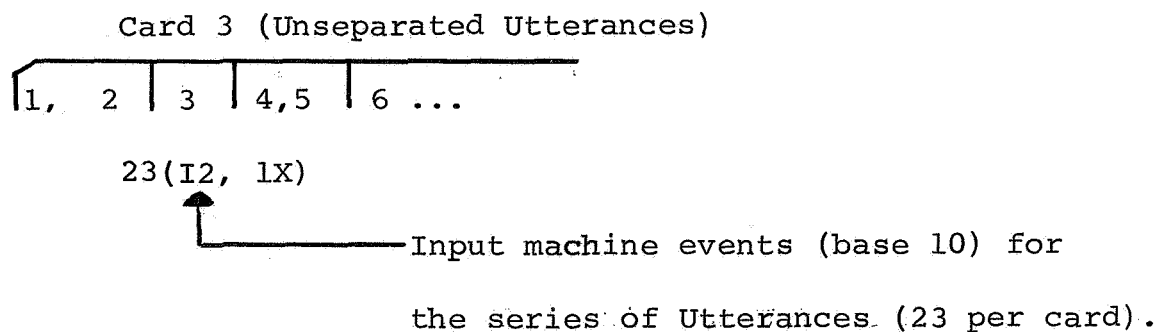
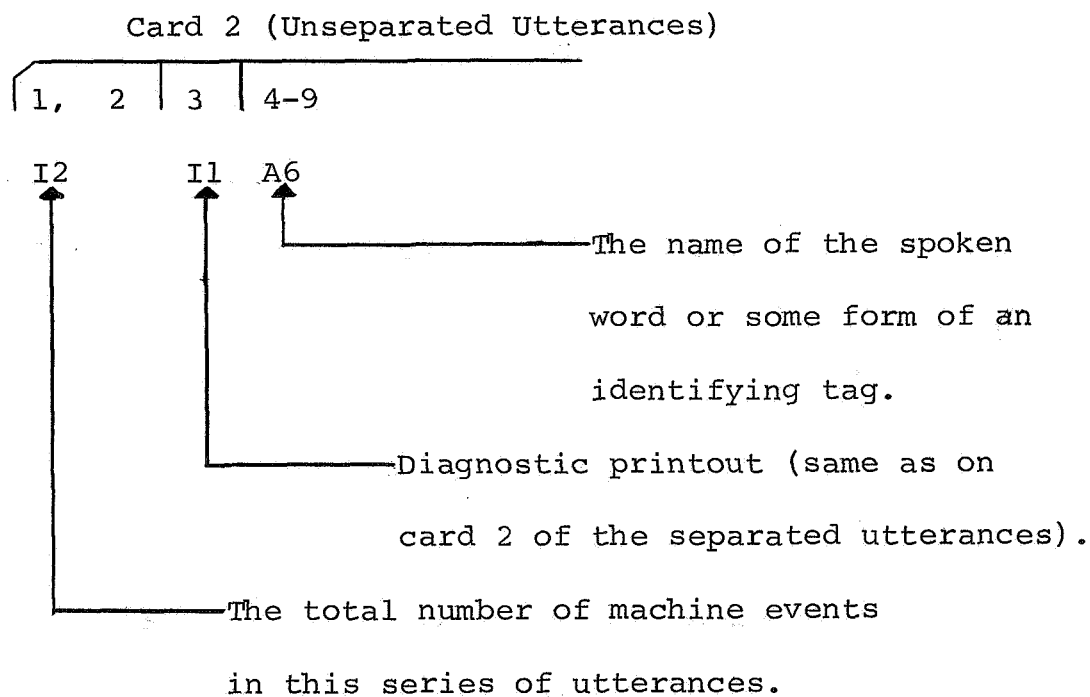
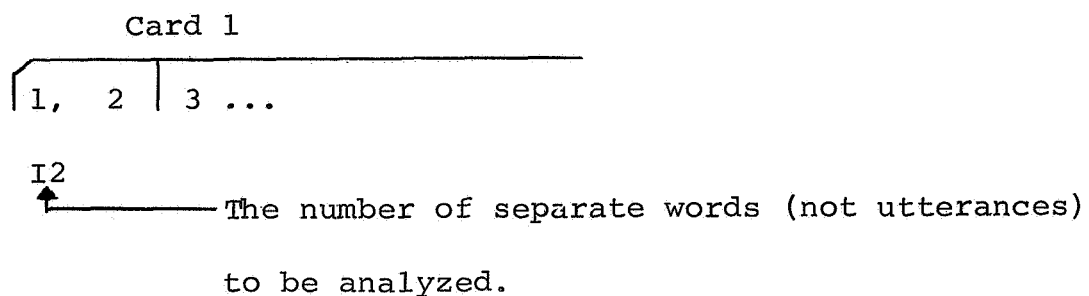
## APPENDIX A

### Input/Output for the Algorithm Program

The input to the program can be of two forms. The first is a series of utterances of a specific word with the start of each utterance specifically identified. The second is a series of utterances with the start of each utterance unspecified. In such a case, the program looks first for events which are represented by a row vector of all zeros. These represent pauses between utterances. As soon as the next non-zero machine event is found, it is considered the start of a new utterance.

The output of the program is of printed form although it can also be in the form of punched cards. The input data are printed out as soon as they are read. Each time the program goes through the algorithm the sequences are printed out along with the significant subsequence found. There is also a diagnostic output. There are two types of diagnostic output. The first lists every comparison made with all the relevant variables. The second lists everything the first lists plus, as indicated, the passage through critical parts of the algorithm along with the associated variables.

The specific input cards are of the following form:



## Card 2 (Separated Utterances)

1,	2	3	4-9	10,11	12	13,14	15 ...
----	---	---	-----	-------	----	-------	--------

I2

I1

A6

23(I2, 1X)

The total number of machine events in each sequence (utterance) 10,11 contains the total for sequence No. 1, 13,14 contains the total for sequence No. 2, etc. Up to 23 separate sequences.

The name of the spoken word in these sequences or an identifying tag of some form.

Diagnostic printout

0 = No printout

1 = Comparison printout

2 = Complete printout

Total number sequences for a particular word (up to 23).

## Card 3 (Separated Utterances)

1,	2	3	4,	5	6	...
----	---	---	----	---	---	-----

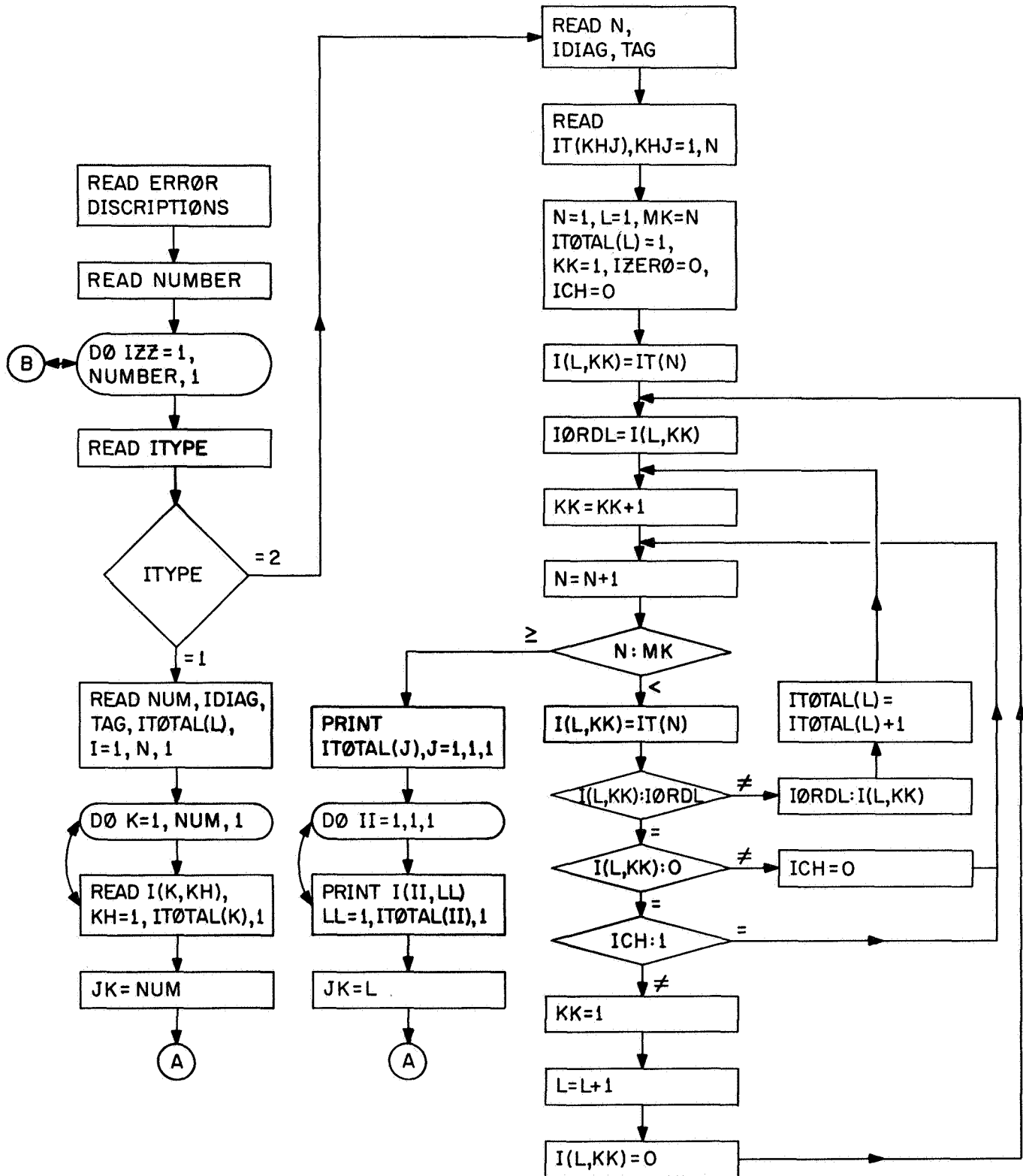
23(I2, 1X)

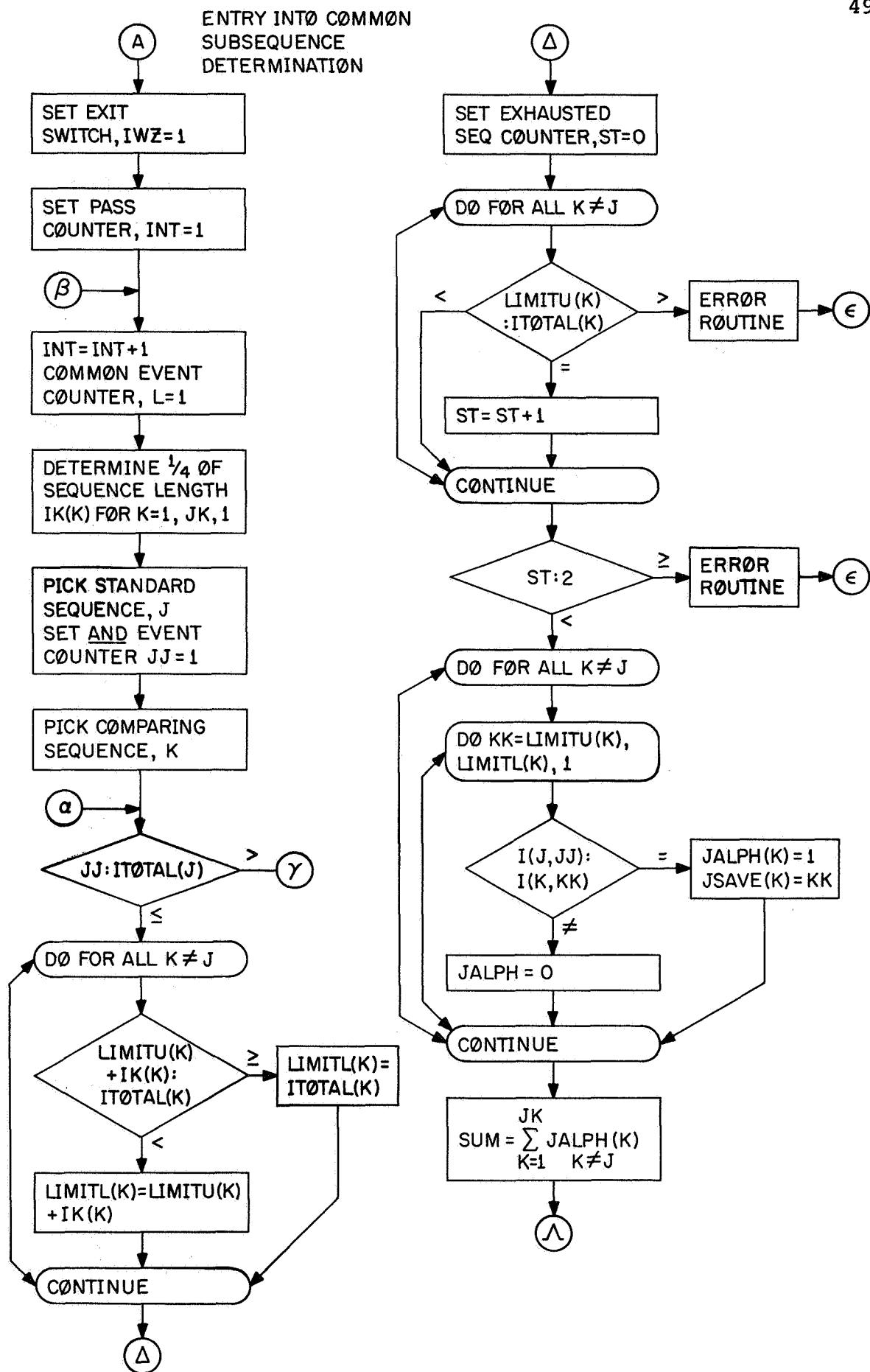


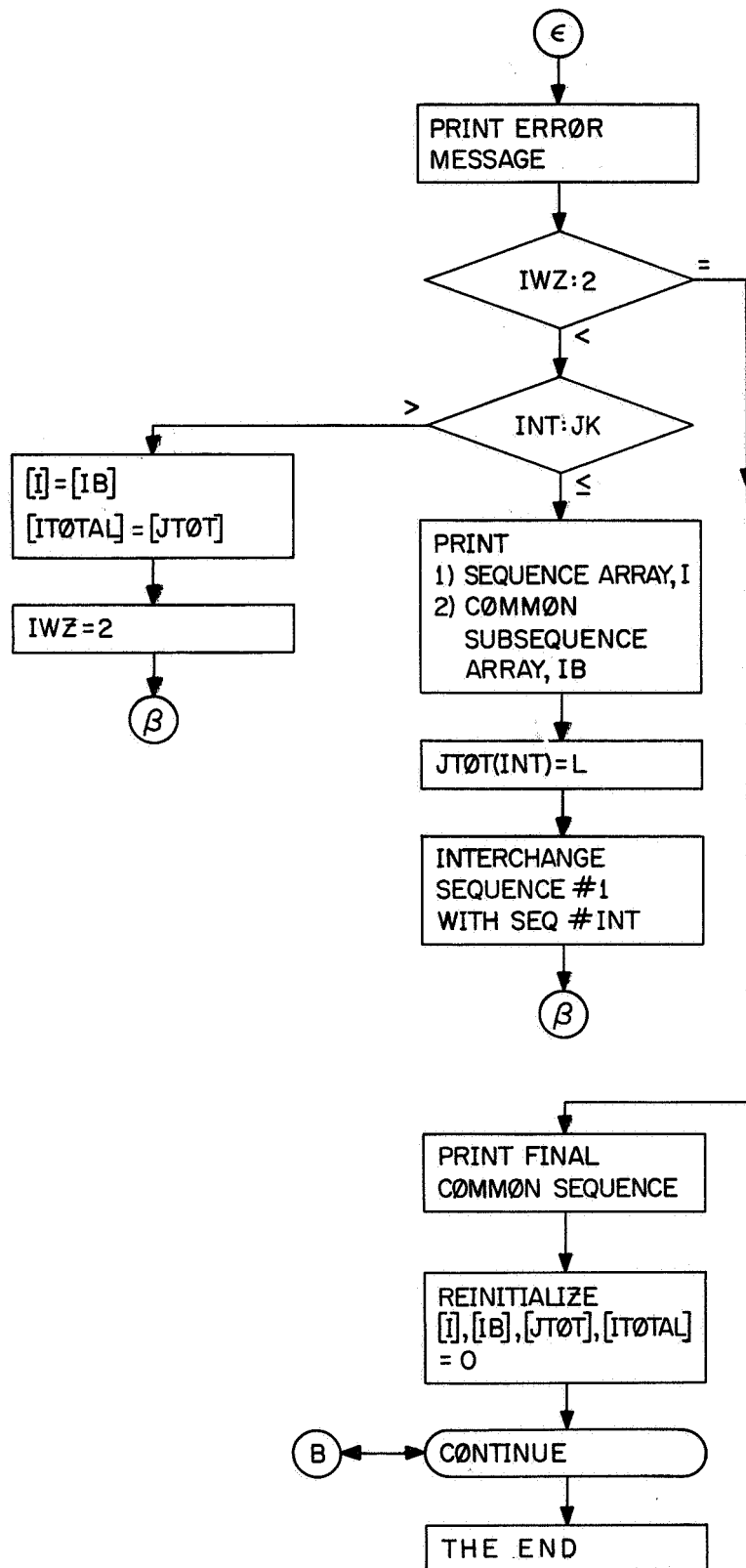
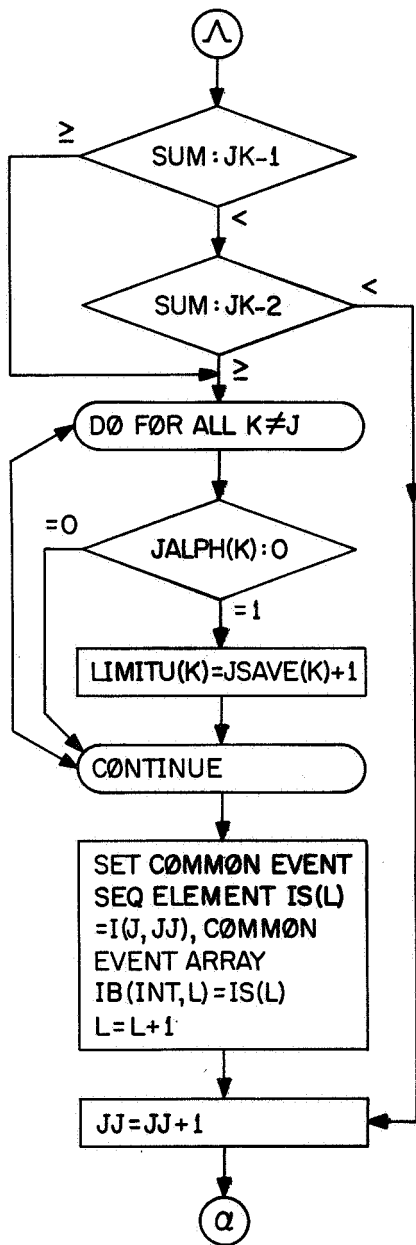
Input machine events (Base 10)

for a sequence up to 23.

The Program Flow Chart









## APPENDIX C

```

$      FORTRAN
$      GO
$      CALL SPRI(1)
C ITOTAL(L)      TOTAL NUMBER OF WORDS IN STRING L      A0000101
C I(L,KK)        WORD NUMBER KK OF STRING L              A0000102
C L              STRING COUNTER                          A0000103
C KK             STRING WORD COUNTER                     A0000104
C IORDL          LAST WORD PROCESSED                     A0000105
C IT(N)          INPUT STRING WORD N                     A0000106
C N              INPUT STRING COUNTER                     A0000107
C XCLO(A,B,C)    EXCLUSIVE OR OF A AND B,C=1 IF TRUE,C=-1 IF FALSE A0000108
C ITEMP          TEMPORARY STORAGE                       A0000109
      N4=25                                              A0000115
      11 FORMAT(23(I2,1X),11X)                          A0000120
      12 FORMAT(1H ,I4,40(1X,I2))
      13 FORMAT(1H1)
      14 FORMAT(1H ,14H STRING NUMBER,1X,I2,35(I2,X1))
      69 FORMAT(1H1)
      99 FORMAT(19H EXCLUSIVE OR ERROR)
      804 FORMAT(1H ,15H INPUT COMPLETE)
      DIMENSION ERROR(10,8)
      DO 569 IJK=1,7,1
      IJL=IJK
      569 RIT 7,568,IJL,(ERROR(IJL,III),III=1,8,1)
      568 FORMAT(I2,8A4)
      DIMENSION JALPH(25),JSAVE(25),LIMITU(25),LIMITL(25)
      DIMENSION IK(25),IB(25,25),JTOT(25)
      DIMENSION I(25,25),ITOTAL(25),M(25),IT(100),IS(40)
      COMMON I
      COMMON ITOTAL
      DIMENSION LOSER(25),TAG(2)
      DIMENSION IZJ(25,6)
C INPUT ROUTINE---(201-229)
      PRINT 69
      READ 3010 ,NUMBER
      PRINT 3010,NUMBER
      PRINT 3333,NUMBER
      3333 FORMAT(1H ,20(I5,1X))
      3010 FORMAT(26(I2,1X))
      DO 3000 IZZ=1,NUMBER,1
      READ 3010,ITYPE
      IF(ITYPE-1) 3001,3001,3002
C SEQUENCES ALREADY SEPARATED
      3001 READ 3011,NUM,IDIAG,TAG(1),TAG(2),(ITOTAL(LJK),LJK=1,NUM,1)
      PRINT 3011,NUM,IDIAG,TAG(1),TAG(2),(ITOTAL(LJK),LJK=1,NUM,1)
      3011 FORMAT(I2,I1,2A3,23(I2,1X))
      DO 3003 IIK=1,NUM,1
      KKK=ITOTAL(IIK)
      READ 11,(I(IIK,KHJ),KHJ=1,KKK,1)
      3003 PRINT 11,(I(IIK,KHJ),KHJ=1,KKK,1)
      JK=NUM
      GO TO 3004
C SEQUENCES NEED TO BE SEPARATED
      3002 READ 3011,N,IDIAG,TAG(1),TAG(2)
      MK=N
      READ 11,(IT(KHJ),KHJ=1,N,1)
      PRINT 11,(IT(KHJ),KHJ=1,N,1)

```

PRINT	804	A2000229
C SEPARATING INPUT STRINGS---	(230-350)	A0000230
ISWIT=1		A2000230
N=1		A0000231
L=1		A0000232
ITOTAL(L)=1		A0000233
KK=1		A0000234
IZERO=0		A0000235
I(L,KK)=IT(N)		A0000240
100 IORDL=I(L,KK)		A0000245
101 KK=KK+1		A0000250
102 N=N+1		A0000255
IF(N-MK)	107,107,90	A0000260
107 I(L,KK)=IT(N)		A0000265
CALL XCLOR(I(L,KK),IORDL,IALPH)		A0000270
1410 IF(IDIAG)	440,440,441	A0000277
441 IKL=1410		A0000278
PRINT	12,IKL,N,L,KK,I(L,KK),IORDL,IALPH,ITOTAL(L),ISWIT	
440 IF(IALPH)	105,104,103	A0000280
105 IORDL=I(L,KK)		A0000285
ITOTAL(L)=ITOTAL(L)+1		A0000290
GO TO	(101,155),ISWIT	
155 ISWIT=1		
GO TO	101	A0000295
103 CALL XCLOR(I(L,KK),IZERO,IALPH)		A0000300
IKL=105		
PRINT	12,IKL,N,L,KK,I(L,KK),IORDL,IALPH,ITOTAL(L),ISWIT	
IF(IALPH)	122,104,106	A0000305
106 ITEMP=I(L,KK)		A0000310
GO TO	(120,102),ISWIT	A0000314
120 I(L,KK)=0		A0000315
ISWIT =2		A0000319
KK=1		A0000320
L=L+1		A0000325
ITOTAL(L)=1		A0000330
I(L,KK)=ITEMP		A0000335
GO TO	100	A0000340
122 ISWIT=1		A0000341
GO TO	105	
104 WOT	6,99	A0000345
GO TO	90	A0000350
C SORTED STRINGS OUTPUT (400-		A0000400
90 CONTINUE		A0000405
PRINT	12,L	A2000410
PRINT	12,(ITOTAL(II),II=1,L,1)	A2000415
DO	91 K=1,L,1	A0000420
LLL=ITOTAL(K)		A0000425
91 PRINT	12,K,(I(K,J),J=1,LLL,1)	A2000430
PRINT	11,(IT(KHJ),KHJ=1,N,1)	A2000433
C LIMITU	THE UPPER LIMIT OF COMPARISION RANGE(THE LOWEST WORD NO.)	A0001000
C LIMITL	THE LOWER LIMIT OF COMPARISION RANGE(THE HIGHEST WD.NO.)	A0001001
C J	SELECTED STRING NUMBER	A0001002
C JJ	SELECTED STRING WORD COUNTER	A0001003
C K	COMPARING STRING COUNTER	A0001004
C KK	COMPARING STRING WORD COUNTER	A0001005
C IWZ	SWITCH TO END AFTER ALL THE REARRANGEMENTS=U,CONT,=2,EXIT	
C M(L)	THE COMMON WORD ARRAY	A0001006

C L	COMMON WORD COUNTER	A0001007
C JK	THE TOTAL NUMBER OF STRINGS	A0001008
C	INTERNAL COMPARE AND VOTE	A0010200
C JALPH(K)	VOTE OF STRING K	A0010201
C JSAVE(K)	SAVED POSITION LOCATION OF SELECTED WORD IN STRING K	A0010202
C JS(L)	SELECTED WORD L	A0010203
C II	VOTE TOTAL	A0010204
C INT	PASS COUNTER	
	JK=L	A0010499
3004	CONTINUE	A0010500
	IWZ=1	A0010501
	PRINT 69	
	INT=0	
3300	CONTINUE	
	INT=INT+1	A0010498
	L=1	A0010501
	DO 900 K=1,JK,1	A0010520
900	LIMITU(K)=1	A0010530
C SET	SEARCH LIMITS	A0010531
	DO 800 K=1,JK,1	A0010532
	ICHECK=8	A0010533
	IK(K)=1.	A0010534
803	IF(ITOTAL(K)-ICHECK) 801,801,802	A0010535
802	IK(K)=IK(K)+1	A0010536
	ICHECK=ICHECK+8	A0010537
	GO TO 803	A0010538
801	IK(K)=IK(K)+1	A0010539
800	CONTINUE	A0010540
	KK=1	A0010545
	K=2	A0010550
	J=1	A0010560
	JJ=1	A0010570
990	IF(JJ-ITOTAL(J)) 999,999,2999	A0010580
999	DO 901 K=2,JK,1	A0010590
902	IF(JJ+IK(K)-ITOTAL(K)) 904,904,903	A0010600
904	LIMITL(K)=JJ+IK(K)	A0010610
	GO TO 901	A0010620
903	LIMITL(K)=ITOTAL(K)	A0010630
901	CONTINUE	A0010640
	ST=0.	A0010650
	DO 996 K=2,JK,1	A0010660
	IF(LIMITU(K)-ITOTAL(K)) 907,908,906	A0010670
906	IJK=5	A0010680
	GO TO 998	A0010690
907	IF(IDIAG ) 400,400,401	A0010697
401	IKL=907	A0010698
	PRINT 12, IKL,J,JJ,I(J,JJ),K,KK,I(K,KK),JK,LIMITU(K),LIMITL(K)	A2010699
400	IF(JJ-IK(K)) 909,912,912	A0010700
	GO TO 102	A0000342
909	IF(LIMITU(K)) 910,912,912	A0010710
910	IJK=6	A0010720
	GO TO 996	A0010730
908	ST=ST+1.	A0010740
912	IF(ST-2.) 916,915,915	A0010750
996	CONTINUE	A0010770
905	K=JK	A0010760
	IF(ST) 916,915,915	A0010780

915 IJK=7	A0010790
GO TO 998	A0010800
916 K=2	A0010810
C THIS DETERMINES IF LIMITU(K) IS TOO LOW AND INITIALIZES LIMITU(2)	A0010999
GO TO 1089	A0011000
1000 CALL XCLOR(I(J,JJ),I(K,KK),IALPH)	A0011010
1400 IF(IDIAG) 430,430,431	A0011017
431 IKL=1000	A0011018
PRINT 12,IKL,J,JJ,I(J,JJ),K,KK,I(K,KK),LIMITU(K),LIMITL(K),JK	A2011019
430 IF(IALPH) 1002,1011,1001	A0011020
1011 IJK=3	A0011030
GO TO 998	A0011040
1001 JALPH(K)=1	A0011050
JSAVE(K)=KK	A0011060
GO TO 1005	A0011070
1002 KK=KK+1	A0011080
1003 IF( IDIAG ) 410,410,411	A0011087
411 IKL=1003	A0011088
PRINT 12, IKL,J,JJ,I(J,JJ),K,KK,I(K,KK),JK,LIMITU(K),LIMITL(K)	A2011089
410 IF(KK-LIMITL(K)) 1000,1000,1004	A0011090
1004 JALPH(K)=0	A0011100
1005 IF(K-JK) 1020,1007,1006	A0011110
1006 IJK=1	A0011120
GO TO 998	A0011130
1100 KK=LIMITU(K)+1	A0011150
GO TO 1000	A0011160
1007 II=0	A0011170
DO 1008 MM=2,JK,1	A0011180
1008 II=II+JALPH(MM)	A0011190
1010 IF(IDIAG) 420,420,421	A0011191
421 IKL=1010	A0011198
PRINT 12, IKL,II	A2011199
420 IF(II-(JK-2)) 1090,1013,1009	A0011200
1009 IF(II-(JK-1)) 1079,1013,1079	A0011201
1079 IJK=2	A0011210
GO TO 998	A0011220
1013 IS(L)=I(J,JJ)	A0011230
IB(INT,L)=I(J,JJ)	
I1=L	
IPIP=I(J,JJ)	
N10=1	
I2=32	
1836 IF(IPIP-I2) 1833,1834,1835	
1835 IZJ(I1,N10)=1	
IPIP=IPIP-I2	
1833 I2=I2/2	
N10=N10+1	
IF(N10-6) 1836,1836,1831	
1834 IZJ(I1,N10)=1	
1831 CONTINUE	
L=L+1	A0011240
DO 1012 N=2,JK,1	A0011250
IF(JALPH(N)) 1014,1012,1015	A0011260
1014 IJK=4	A0011270
GO TO 998	A0011280
1015 LIMITU(N)=JSAVE(N)	A0011290
1012 CONTINUE	A0011300

1090 JJ=JJ+1	A0011310
GO TO 990	A0011320
1020 K=K+1	A0011330
1089 IF(JJ-1) 1101,1101,1100	A0011340
1101 KK=LIMITU(K)	A0011350
GO TO 1000	A0011360
998 PRINT 997,IJK,(ERROR(IJK,III),III=1,8,1),J,JJ,K,KK,I(J,JJ),	A2012100
1 L	A2012110
997 FORMAT(1H ,I2,2X,8A4,70(2X,I3))	A0012112
C REMEMBER TO READ IN ERROR MESSAGES	A0012113
2999 PRINT 3015,TAG(1),TAG(2)	A2012114
L=L-1	
3015 FORMAT(1H ,14H FOR THE WORD,2A3)	A0012115
DO 3039 J=1,JK,1	A0012116
II=ITOTAL(J)	A0012117
3039 PRINT 12,(I(J,K),K=1,II,1)	
PRINT 2002,(IB(INT,MJ),MJ=1,L,1)	
2002 FORMAT(1H ,33H THE COMMON SEQUENCE OF EVENTS IS,2X,30(I2,1X)/	A0012120
120(I2,1X))	A0012121
PRINT 1837, ((IZJ(J1,J2),J2=1,6,1),IB(INT,J1),J1=1,L,1)	
1837 FORMAT(1H ,6I1,2X,I3)	
JTOT(INT)=L	A0012130
DO 1838 J1=1,25,1	
DO 1838 J2=1,6,1	
1838 IZJ(J1,J2)=0	
C INTERCHANGE OF STRINGS TO IMPROVE PREDICTION	A0012210
PRINT 12,(IB(INT,K),K=1,L,1)	2
IF(IDIAG) 432,432,433	
433 IKL=880	
PRINT 12,IKL,IWZ,INT	2
432 IF(IWZ-2) 851,3030,3030	A0012211
850 IF(INT-JK) 840,840,860	A0012220
840 N2=N2+1	A0013100
CALL CHANGE(N2,N3,N4)	A0013110
CALL CHANGE(1,N2,N4)	A0013120
N3=N2	A0013130
GO TO 3300	A0013139
851 IF(INT-1) 830,830,850	
C FIRST PASS INTERCHANGE	
830 N1=1	A0013200
N2=2	A0013210
N3=N2	A0013220
CALL CHANGE(1,N2,25)	A0013230
GO TO 3300	A0013240
C CHECK STRINGS	A0014000
860 PRINT 881	A2014010
881 FORMAT(1H ,16H COMMON STRINGS FOUND)	A0014020
DO 882 J=1,JK,1	A0014030
PRINT 12,J,(IB(J,K),K=1,N4,1)	A2014040
ITOTAL(J)=JTOT(J)	A0014050
DO 882 K=1,N4,1	A0014060
882 I(J,K)=IB(J,K)	A0014070
C SET SWITCH FOR EXIT ON NEXT PASS	
IWZ=2	A0014080
GO TO 3300	
3030 CONTINUE	
C INITIALIZE THE MATRICES	A0120010

DO 3040 J=1,25,1	A0120020
IK(J)=0	A0120021
JTOT(J)=0	A0120022
IS(J)=0	A0120030
DO 3040 K=1,25,1	A0120040
I(J,K)=0	
3040 IB(J,K)=0	
PRINT 69	
3000 CONTINUE	A0123000
2001 FORMAT(1H1,15H END OF PROGRAM)	A0123222
2000 PRINT 2001	A2123221
END	A0123223
\$ FORTRAN	
\$ PUNCH OBJECT	
C SWITCH ROUTINE	CHANG000
SUBROUTINE CHANGE(N1,N2,N3)	CHANG001
DIMENSION I(25,25),ITOTAL(25)	
COMMON I	
COMMON ITOTAL	
811 DO 810 K=1,N3,1	CHANG012
IEMP=I(N1,K)	CHANG013
I(N1,K)=I(N2,K)	CHANG014
810 I(N2,K)=IEMP	CHANG015
IEMP=ITOTAL(N1)	
ITOTAL(N1)=ITOTAL(N2)	
ITOTAL(N2)=IEMP	
RETURN	CHANG016
END	CHANG017
\$ FORTRAN	
SUBROUTINE XCLOP (IORD1,IORD2,IALPH)	
IF (IORD1-IORD2) 1,2,1	
2 IALPH=1	
GO TO 3	
1 IALPH=-1	
3 CONTINUE	
RETURN	
END	

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1. ORIGINATING ACTIVITY <i>(Corporate author)</i> University of Illinois Department of Electrical Engineering Urbana, Illinois 61801		2a. REPORT SECURITY CLASSIFICATION UNCLASSIFIED
		2b. GROUP
3. REPORT TITLE AUTOMATIC DETERMINATION OF INVARIANCE IN MACHINE CODED SPEECH		
4. DESCRIPTIVE NOTES <i>(Type of report and inclusive dates)</i> Scientific; ; interim		
5. AUTHOR(S) <i>(Last name, first name, initial)</i>  Schill, John		
6. REPORT DATE December 1, 1968	7a. TOTAL NO. OF PAGES 56	7b. NO. OF REFS 15
8a. CONTRACT OR GRANT NO. AF-AFOSR 7-67	9a. ORIGINATOR'S REPORT NUMBER(S) BCL Report 7.4	
b. PROJECT AND TASK NO.  c. 9769-04 d. 61445014 681304	9b. OTHER REPORT NO(S) <i>(Any other numbers that may be assigned this report)</i>	
10. AVAILABILITY/LIMITATION NOTICES  Distribution of this document is unlimited.		
11. SUPPLEMENTARY NOTES Partial sponsorship NASA NGR 14-005-111 and AF 33(615)-3890		12. SPONSORING MILITARY ACTIVITY Directorate of Information Sciences Air Force Office of Scientific Research Arlington, Virginia 22209
13. ABSTRACT  A method for determining significant patterns for identification of spoken words was developed. This method is based on the work of Gazdag using his concept of significant subsequences (SSS) which is based upon a machine representation of spoken words. A system was assembled using an algorithm for ascertaining these SSS's implemented on the ILLIAC II. This system consisted of an encoder, a sampler, for the encoder and digital computer. The spoken digits were recognized by this system using the concept of the SSS.		

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	ROLE	WT	ROLE	WT	ROLE	WT
speech recognition algorithm						

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